# Real-Time Audio Translator - Setup & Usage Guide

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### Installation

### **Prerequisites**

- Python 3.8 or higher
- pip (Python package manager)
- Git (for cloning the repository)

#### **Step 1: Clone the Repository**

bash

git clone https://github.com/yourusername/real-time-translator.git

cd real-time-translator

### **Step 2: Create Virtual Environment (Recommended)**

bash

# Windows

python -m venv venv venv\Scripts\activate

# macOS/Linux

python3 -m venv venv

source venv/bin/activate

### **Step 3: Install Dependencies**

### **Option A: Standard Installation**

bash

pip install -r requirements.txt

#### **Option B: GPU Support (Faster processing)**

bash

# For NVIDIA GPUs with CUDA 11.8

pip install torch torchvision torchaudio --index-url https://download.pytorch.org/whl/cu118 pip install -r requirements.txt

### **Step 4: Verify Installation**

bash

python main\_entry.py --check-deps

### **Ø** Initial Setup

#### 1. First Launch

bash

python main\_entry.py

### 2. Configure Audio Device

#### **For Windows Users:**

- 1. Physical Microphone: Select your microphone from the device list
- 2. System Audio:
  - Install <u>VB-Audio Virtual Cable</u>
  - Set virtual cable as Windows default playback device
  - Select "CABLE Output" in the translator

#### For macOS Users:

- 1. Physical Microphone: Select from device list
- 2. System Audio:
  - Install BlackHole

- Create Multi-Output Device in Audio MIDI Setup
- Select BlackHole in the translator

#### **For Linux Users:**

1. Use PulseAudio loopback:

bash
pactl load-module module-loopback

2. Select "Monitor" device in the application

#### 3. Test Your Setup

- 1. Click " Refresh" to scan devices
- 2. Select your audio source
- 3. Click " Test Device"
- 4. Play audio or speak you should see the level meter move
- 5. Adjust threshold based on the test results

#### 4. Load Al Models

- 1. Click " Load Al Models"
- 2. First-time loading downloads models (~500MB-1GB)
- 3. Wait for " Models Loaded" confirmation
- 4. Models are cached for future use

### Using the Application

#### **Basic Workflow**

#### 1. Start Translation

- Ensure models are loaded
- Select source language (or use Auto-detect)
- Select target language
- Click " Start Translation"

#### 2. During Translation

• Green level meter shows audio activity

- Original text appears in blue
- Translations appear in green
- Timestamps show when each segment was processed

#### 3. Stop Translation

- Click " Stop Translation"
- Session statistics are displayed

### **Understanding the Interface**

#### **Audio Configuration Panel**

• Audio Device: Source for audio capture

• Audio Level: Real-time volume indicator

Speech Threshold: Minimum level to trigger processing

• Audio Gain: Amplification factor

#### **Language Configuration Panel**

• Source Language: Language being spoken

• Target Language: Language to translate to

• Al Model: Whisper model size (accuracy vs. speed)

### **Output Display Controls**

• **Show Logs**: Toggle system messages

• **Show Timestamp**: Toggle time stamps

• Show Process Time: Toggle performance metrics

• Show Original: Toggle source text display

#### **Common Use Cases**

### 1. Watching Foreign Videos

- 1. Set system audio as default playback
  - 2. Select video's language as source
  - 3. Select your language as target
  - 4. Start translation and play video

#### 2. Online Meetings

- 1. Use virtual audio cable
- 2. Route meeting audio through cable
- 3. Set source to "Auto-detect"
- 4. Start translation before meeting

#### 3. Language Practice

- 1. Use microphone input
- 2. Set source to language you're practicing
- 3. Set target to your native language
- 4. Speak and see instant feedback



### Advanced Configuration

### **Optimizing Performance**

#### **For Accuracy**

```
json
 "whisper_model": "medium",
 "energy_threshold": 0.005,
 "min_speech_duration": 0.5,
 "silence_timeout": 1.0
}
```

#### **For Speed**

```
json
 "whisper_model": "tiny",
 "energy_threshold": 0.02,
 "min_speech_duration": 0.3,
 "silence_timeout": 0.5
```

### **Custom Language Models**

To add new translation pairs:

- 1. Edit | constants.py
- 2. Add model mapping to (TRANSLATION\_MODELS)
- 3. Example:

```
python
("ko", "en"): "Helsinki-NLP/opus-mt-ko-en"
```

### **Adjusting Audio Settings**

#### **For Noisy Environments**

- Increase threshold to 0.02-0.05
- Decrease gain to 0.5-0.8
- Use noise suppression software

#### **For Quiet Sources**

- Decrease threshold to 0.001-0.005
- Increase gain to 2.0-3.0
- Position closer to audio source

### **Propriet** Tips & Best Practices

### **Audio Quality**

- 1. **Use high-quality audio sources** Clear audio improves accuracy
- 2. Minimize background noise Use noise cancellation when possible
- 3. Maintain consistent volume Avoid sudden volume changes

### **Performance Optimization**

- 1. Close unnecessary programs Free up CPU/RAM
- 2. **Use GPU if available** 3-5x faster processing
- Start with smaller models Upgrade if needed

### **Language Tips**

- 1. Auto-detect works best with clear speech
- 2. **Specify language when possible** Faster and more accurate
- 3. Similar languages may need manual selection

### **Troubleshooting Quick Fixes**

Problem	Solution	
No audio detected	Check device selection and permissions	
Delayed translations	Use smaller model or upgrade hardware	
Wrong language detected	Manually select source language	
Poor accuracy	Use larger model, improve audio quality	
Application crashes	Check RAM usage, restart application	
<b>▲</b>	•	

## **Keyboard Shortcuts (Future Feature)**

- Space: Pause/Resume translation
- (Ctrl+C): Copy selected text
- (Ctrl+S): Save translation history
- (Esc): Stop translation

# **III** Performance Expectations

### **Model Comparison**

Model	Speed	Accuracy	RAM Usage	Best For
Tiny	44444	☆☆	~1GB	Real-time, clear audio
Base	4444	<b>☆☆☆</b>	~1.5GB	Balanced performance
Small	444	<b>☆☆☆☆</b>	~2GB	Better accuracy
Medium	44	***	~5GB	Professional use

### **System Requirements by Use Case**

Use Case	Min RAM	Recommended	GPU Benefit
Personal	4GB	8GB	Optional
Professional	8GB	16GB	Recommended
Streaming	8GB	16GB	Required
4	•	•	<b>&gt;</b>

# Getting Help

#### Resources

- **Documentation**: Check project README
- **Issues**: GitHub issue tracker
- Community: CS50 Discord

### **Debug Mode**

Run with verbose output:

bash python main\_entry.py --debug

### **Logs Location**

- Windows: (%APPDATA%\RealTimeTranslator\logs)
- macOS: (~/Library/Logs/RealTimeTranslator)
- Linux: (~/.local/share/RealTimeTranslator/logs)

Happy Translating! 🔵 🏂

